

## ABSTRACT OF THE DISCLOSURE

A wraparound delay amount detecting part calculates a cross-correlation  $r(k)$  from an output speech signal “ $a_i$ ” supplied to a loudspeaker and an input speech signal “ $b_i$ ” inputted through a microphone array to obtain a delay amount “ $d$ ” of a wraparound speech signal. The delay processing part generates a speech signal “ $a_{i-d}$ ” obtained by delaying the output speech signal “ $a_i$ ” by the delay amount “ $d$ ”. Even if there is a change in delay amount due to the variation in environment, appropriate delay processing can be conducted by the delay processing part. In an adaptive filter, an estimated wraparound speech signal  $a_{i-d}'$  is generated from the speech signal “ $a_{i-d}$ ” subjected to delay processing. A subtracter subtracts the estimated wraparound speech signal  $a_{i-d}'$  from the input speech signal “ $b_i$ ” to generate an echo cancellation signal “ $e_i$ ”. A coefficient updating part updates the coefficient of the adaptive filter.